Ref #	Hits	Search Query	DBs	Default Operator	Plurals	Time Stamp
S1	1	audio.ab. and (audio adj sampl\$4) and (filter\$4 same dc) and (autocorrelation "auto-correlation") and microphon\$3 and coefficient and ((real adj time) "real-time")	US-PGPUB; USPAT; EPO; JPO; DERWENT; IBM_TDB	OR	ON	2005/03/04 06:37
S2	1	audio.ab. and (audio adj sampl\$4) and (filter\$4 same dc) and (autocorrelation "auto-correlation") and microphon\$3 and coefficient and ((real adj time) "real-time" realtime)	US-PGPUB; USPAT; EPO; JPO; DERWENT; IBM_TDB	OR	ON	2005/03/04 06:38
S3 ·	1	audio.ab. and (audio adj sampl\$4) and (filter\$4 same dc) and (autocorrelation "auto-correlation") and microphon\$3.ab. and coefficient and ((real adj time) "real-time" realtime)	US-PGPUB; USPAT; EPO; JPO; DERWENT; IBM_TDB	OR	ON	2005/03/04 06:38
S4	3029	audio adj sample	US-PGPUB; USPAT; EPO; JPO; DERWENT; IBM_TDB	OR	ON	2005/03/04 06:38
S5	3598	audio adj sampl\$3	US-PGPUB; USPAT;	OR	ON	2005/03/04 06:38
			EPO; JPO; DERWENT; IBM_TDB		, , , ,	
S6	1940	audio.ab. and (audio adj sampl\$3)	US-PGPUB; USPAT; EPO; JPO; DERWENT; IBM_TDB	OR	ON	2005/03/04 06:38
S7	279	audio.ab. and (audio adj sampl\$3) and (microphone same (connect\$4 detect\$4))	US-PGPUB; USPAT; EPO; JPO; DERWENT; IBM_TDB	OR	ON	2005/03/04 06:39
S8	163	audio.ab. and (audio adj sampl\$3) and (microphone same (connect\$4 detect\$4))	USPAT	OR	ON	2005/03/04 06:40
S9	19	audio.ab. and (audio adj sampl\$3) same (microphone same (connect\$4 detect\$4))	USPAT	OR	ON	2005/03/04 06:40
S10	1542518	speech near\$7 detect\$4	USPAT	OR	ON	2005/03/04 06:40
S11	139	S7 and S10	USPAT	OR	ON .	2005/03/04 06:40
S12	169904	S10 and coefficient	USPAT	OR	ON	2005/03/04 06:41

S13	169904	S10 and coefficient	US-PGPUB; USPAT; EPO; JPO; DERWENT; IBM_TDB	OR	ON	2005/03/04 06:42
S14	21	S10 and (autocorrelation "auto-correlation") and microphon\$3.ab. and coefficient and ((real adj time) "real-time" realtime)	US-PGPUB; USPAT; EPO; JPO; DERWENT; IBM_TDB	OR	ON	2005/03/04 06:43
S15		"l14" and speech.ab.	US-PGPUB; USPAT; EPO; JPO; DERWENT; IBM_TDB	OR	ON	2005/03/04 06:45
S16	40	microphone.clm. and audio and (autocorrelation "auto-correlation") and coefficient and ((real adj time) "real-time" realtime)	US-PGPUB; USPAT; EPO; JPO; DERWENT; IBM_TDB	OR	ON	2005/03/04 06:59
S17	13	"5825771"	US-PGPUB; USPAT; EPO; JPO; DERWENT; IBM_TDB	OR	ON	2005/03/04 06:59

'n

IEEE HOME | SEARCH IEEE | SHOP | WEB ACCOUNT | CONTACT IEEE



Membership Publications/Services Standards Conferences IEEE Xolo 1 Million Doci Welcome 1 Million User United States Patent and Trademark Office And Grov » Search Res FAQ Terms IEEE Peer Review **Quick Links** Ξ Welcome to IEEE Xplore® Your search matched 22 of 1131693 documents. C Home A maximum of **500** results are displayed, **15** to a page, sorted by **Relevance** in — What Can I Access? **Descending** order. C Log-out Refine This Search: **Tables of Contents** You may refine your search by editing the current search expression or entering a new one in the text box. — Journals & Magazines (autocorrelation <or> auto correlation) <and> (audio < Search Conference Check to <u>search</u> within this result set **Proceedings** ()- Standards Results Key: JNL = Journal or Magazine CNF = Conference STD = Standard Search O- By Author 1 All-pole modeling technique based on weighted sum of LSP polynomials O- Basic Backstrom, T.; Alku, P.; — Advanced Signal Processing Letters, IEEE, Volume: 10, Issue: 6, June 2003 O- CrossRef Pages: 180 - 183 Member Services [PDF Full-Text (277 KB)] [Abstract] **IEEE JNL** ()- Join IEEE 2 Speech analysis and segmentation by parametric filtering C - Establish IEEE Ta-Hsin Li; Gibson, J.D.; Web Account Speech and Audio Processing, IEEE Transactions on , Volume: 4 , Issue: 3 , May ()- Access the 1996 **IEEE Member** Pages: 203 - 213 Digital Library **IEEE Enterprise** [Abstract] [PDF Full-Text (1116 KB)] **IEEE JNL** ()- Access the 3 Noise cancellation for hearing aids **IEEE Enterprise** Chazan, D.; Medan, Y.; Shvadron, U.; File Cabinet Acoustics, Speech, and Signal Processing [see also IEEE Transactions on Signal Processing], IEEE Transactions on , Volume: 36 , Issue: 11 , Nov. 1988 Print Format Pages:1697 - 1705 [Abstract] [PDF Full-Text (864 KB)] 4 Multiple pitch estimation of poly-phonic audio signals in a frequency-lag domain using the bispectrum Abeysekera, S.S.; Circuits and Systems, 2004. ISCAS '04. Proceedings of the 2004 International Symposium on , Volume: 3 , 23-26 May 2004

IEEE CNF

[PDF Full-Text (336 KB)]

Pages: III - 469-72 Vol. 3

[Abstract]

5 Implementation of a real-time <u>audio</u> watermark extractor on ARM940T

Mathew, M.; Yoonhark Oh; Jaeyoung Lee; Changhoon Yim; Kilsu Eo; Signals, Circuits and Systems, 2003. SCS 2003. International Symposium on , Volume: 2 , 10-11 July 2003

Pages:605 - 608 vol.2

[Abstract] [PDF Full-Text (347 KB)] IEEE CNF

6 A linear predictive method using extrapolated samples for modelling of voiced speech

Varho, S.; Alku, P.;

Applications of Signal Processing to $\underline{\text{Audio}}$ and Acoustics, 1997. 1997 IEEE ASSP Workshop on , 19-22 Oct. 1997

Pages:4 pp.

[Abstract] [PDF Full-Text (420 KB)] IEEE CNF

7 An extended multidelay filter: fast low-delay algorithms for very highorder adaptive systems

Buchner, H.; Kellermann, W.; Benesty, J.; Acoustics, Speech, and Signal Processing, 2003. Proceedings. (ICASSP '03). 2003 IEEE International Conference on , Volume: 5 , 6-10 April 2003 Pages:V - 385-8 vol.5

[Abstract] [PDF Full-Text (322 KB)] IEEE CNF

8 All-pole modeling technique based on the weighted sum of the LSP polynomials

Alku, P.; Backstrom, T.;

Acoustics, Speech, and Signal Processing, 2002. Proceedings. (ICASSP '02). IEEE International Conference on , Volume: 1 , 13-17 May 2002

Pages:I-665 - I-668 vol.1

[Abstract] [PDF Full-Text (333 KB)] IEEE CNF

9 Efficient calculation of a physiologically-motivated representation for sound

Klapuri, A.P.; Astola, J.T.;

Digital Signal Processing, 2002. DSP 2002. 2002 14th International Conference

on , Volume: 2 , 1-3 July 2002

Pages: 587 - 590 vol.2

[Abstract] [PDF Full-Text (366 KB)] IEEE CNF

10 A microphone array with adaptive post-filtering for noise reduction in reverberant rooms

Zelinski, R.;

Acoustics, Speech, and Signal Processing, 1988. ICASSP-88., 1988 International Conference on , 11-14 April 1988

Pages: 2578 - 2581 vol. 5

[Abstract] [PDF Full-Text (340 KB)] IEEE CNF

11 Time domain design of nonrecursive least mean-square digital filters Kellogg, W.;

Audio and Electroacoustics, IEEE Transactions on , Volume: 20 , Issue: 2 , Jun

1972

Pages: 155 - 158

[Abstract] [PDF Full-Text (344 KB)] TEEE JNL

12 Spectral analysis of the call of the male killer whale

Singleton, R.; Poulter, T.;

Audio and Electroacoustics, IEEE Transactions on , Volume: 15 , Issue: 2 , Jun

1967

Pages: 104 - 113

[PDF Full-Text (1448 KB)] [Abstract] **IEEE JNL**

13 New methods of pitch extraction

Sondhi, M.;

Audio and Electroacoustics, IEEE Transactions on , Volume: 16 , Issue: 2 , Jun

1968

Pages: 262 - 266

[Abstract] [PDF Full-Text (464 KB)] **IEEE JNL**

14 The SIFT algorithm for fundamental frequency estimation

Markel, J.;

Audio and Electroacoustics, IEEE Transactions on , Volume: 20 , Issue: 5 , Dec

1972

Pages: 367 - 377

[Abstract] [PDF Full-Text (1176 KB)] **IEEE JNL**

15 A comparative performance study of several pitch detection algorithms

Rabiner, L.; Cheng, M.; Rosenberg, A.; McGonegal, C.;

Acoustics, Speech, and Signal Processing [see also IEEE Transactions on Signal Processing], IEEE Transactions on , Volume: 24 , Issue: 5 , Oct 1976

Pages: 399 - 418

[Abstract] [PDF Full-Text (1880 KB)] **IEEE JNL**

1 2 Next

Home | Log-out | Journals | Conference Proceedings | Standards | Search by Author | Basic Search | Advanced Search | Join IEEE | Web Account | New this week | OPAC Linking Information | Your Feedback | Technical Support | Email Alerting | No Robots Please | Release Notes | IEEE Online Publications | Help | FAQ Terms Back to Top

Copyright © 2004 IEEE - All rights reserved



<u>Subscribe</u> (Full Service) <u>Register</u> (Limited Service, <u>Free</u>) <u>Login</u>

Search: • The ACM Digital Library • C The Guide

(autocorrelation "auto-correlation") and microphone and coeffi

SEARCH

YAMARUJ JAHIBUOLISA SHIT

Feedback Report a problem Satisfaction survey

Terms used <u>autocorrelation</u> <u>auto</u> <u>correlation</u> and <u>microphone</u> and <u>coefficient</u> and <u>real time</u> <u>real</u> <u>time</u> <u>realtime</u>

Found 17,607 of 151,219

Sort results by relevance

Display results expanded form

Save results to a Binder

Search Tips

Try an <u>Advanced Search</u>
Try this <u>search</u> in <u>The ACM Guide</u>

Open results in a new window

Results 1 - 20 of 200.

Best 200 shown

Result page: 1 2 3 4 5 6 7 8 9 10 next

Polovon

Relevance scale 🗆 📟 📰 🔳

1 An adaptive congestion control scheme for real-time packet video transport Hemant Kanakia, Partho P. Mishra, Amy Reibman

October 1993 ACM SIGCOMM <u>Computer</u> Communication Review , Conference proceedings on Communications architectures, protocols and applications, Volume 23 Issue 4

Full text available: pdf(1.26 MB)

Additional Information: full citation, references, citings, index terms

2 Cloth & deformable bodies: Feel the "fabric": an audio-haptic interface

G. Huang, D. Metaxas, M. Govindaraj

July 2003 Proceedings of the 2003 ACM SIGGRAPH/Eurographics Symposium on Computer animation

Full text available: pdf(6.08 MB)

Additional Information: full citation, abstract, references, index terms

An objective fabric modeling system should convey not only the visual but also the haptic and <u>audio</u> sensory feedbacks to remote/internet users via an audio-haptic interface. In this paper we develop a fabric surface <u>property</u> modeling system consisting of a stylus based fabric characteristic sound modeling, and an audio-haptic interface. By using a stylus, <u>people</u> can perceive fabrics surface roughness, friction, and softness though not as precisely as with their bare fingers. The audio-haptic int ...

3 <u>Authentication II: Audio watermarking algorithm for real-time speech integrity and authentication</u>

Song Yuan, Sorin A. Huss

September 2004 Proceedings of the 2004 <u>multimedia</u> and <u>security</u> workshop on <u>Multimedia</u> and <u>security</u>

Full text available: pdf(259.52 KB)

Additional Information: full citation, abstract, references, index terms

Data integrity and source origin authentication are essential topics for real-time <u>multimedia</u> systems. But traditional method, such as MAC, is not very applicable to overcome the distortion introduced in real-time <u>multimedia</u> communication. In this paper a new integrity mechanics deploying speech watermarking is presented. The advocated approach adopts public key encryption to efficiently generate non-repudiate speech. In the last part of the article, a speech watermarking algorithm incorporating ...

Keywords: integrity and source origin authentication, real-time <u>multimedia</u> communication and <u>internet</u> telephony, speech watermarking

4 System architectures for computer music

	·				
	John W. Gordon June 1985 ACM Computing Surveys (CSUR), Volume 17 Issue 2				
	Full text available: pdf(4.61 MB) Additional Information: full citation, abstract, references, citings, index terms, review				
	Computer music is a relatively new field. While a large proportion of the public is aware of computer music in one form or another, there seems to be a need for a better understanding of its capabilities and limitations in terms of synthesis, performance, and recording hardware. This article addresses that need by surveying and discussing the architecture of existing computer music systems. System requirements vary according to what the system will be used for. Common uses for co				
5	Special issue on independent components analysis: A maximum likelihood approach to				
	single-channel source separation Gil-Jin Jang, Te-Won Lee				
December 2003 The Journal of Machine Learning Research, Volume 4					
	Full text available: pdf(1.07 MB) Additional Information: full citation, abstract, index terms				
	This paper presents a new technique for achieving blind signal separation when given only a single channel recording. The main concept is based on exploiting a priori sets of time-domain basis functions learned by independent component analysis (ICA) to the separation of mixed source signals observed in a single channel. The inherent time structure of sound sources is reflected in the ICA basis functions, which encode the sources in a statistically efficient manner. We derive a le				
6	Session 2: modeling: A flow-based model for internet backbone traffic				
Chadi Barakat, Patrick Thiran, Gianluca Iannaccone, Christophe Diot, Philippe Owezar November 2002 Proceedings of the 2nd ACM SIGCOMM Workshop on Internet measurment					
	Full text available: pdf(1.35 MB) Additional Information: full citation, abstract, references, citings, index terms				
	Our goal is to <u>design</u> a <u>traffic</u> model for uncongested IP backbone links that is simple enough to be used in <u>network</u> operation, and that is protocol and application agnostic in order to be as general as possible. The proposed solution is to model the <u>traffic</u> at the flow level by a Poisson shot-noise process. In our model, a flow is a generic notion that must be able to capture the characteristics of any kind of data stream. We analyze the accuracy of the model with real <u>traffic</u> traces collected o				
7	Automating road surface analysis				
	L. Donnell Payne March 1992 Proceedings of the 1992 ACM/SIGAPP symposium on Applied computing:				
	technological challenges of the 1990's Full text available: pdf(720.06 KB) Additional Information: full citation, references, index terms				
	Tuli text available. (A. pai(720.00 Kb)				
8	Invited paper: In-situ speech visualization in real-time interactive installation and				
	<u>performance</u> Golan Levin, Zachary Lieberman				
	June 2004 Proceedings of the 3rd international symposium on Non-photorealistic animation and rendering				
	Full text available: pdf(71.28 MB) Additional Information: full citation, abstract, references				
	Although we can sense someone's vocalizations with our ears, nose, and haptic sense, speech is invisible to us without the help of technical aids. In this paper, we present three interactive artworks which explore the question: "if we could see our speech, what might it look like?" The artworks we present are concerned with the aesthetic implications of making the human voice visible, and were created with a particular emphasis on interaction designs that support the				

perception of tight spatio-t ...

Keywords: art, audiovisual performance, augmented reality, computer vision, head tracking, interactive installation, phonesthesia, sound-image relationships, speech analysis, speech visualization

9 Emerging technologies: WLANs and WPANs: Markov-based modeling of wireless local area networks

Syed A. Khayam, Hayder Radha

September 2003 Proceedings of the 6th ACM international workshop on Modeling analysis and simulation of wireless and mobile systems

Full text available: pdf(317.87 KB) Additional Information: full citation, abstract, references, index terms

Errors introduced by a wireless medium are more frequent and profound than contemporary wired media. Some of these errors, which are not corrected by the physical layer, result in Medium Access Control (MAC) layer bit errors and packet losses. Design of wireless protocols and applications can benefit substantially from a thorough understanding of these MAC layer impairments. This paper evaluates and proposes Markov-based stochastic chains to model the 802.11b MAC-to-MAC channel behavior for both ...

Keywords: 802.11b networks, MAC-to-MAC, Markov-based modeling, information theoretic evaluation

10 Continuous media communication with dynamic QOS control using ARTS with an FDDI network

Hideyuki Tokuda, Yoshito Tobe, Stephen T.-C. Chou, José M. F. Moura

October 1992 ACM SIGCOMM Computer Communication Review , Conference proceedings on Communications architectures & protocols, Volume 22 Issue 4

Full text available: pdf(1.04 MB)

Additional Information: full citation, abstract, references, citings, index terms

Continuous media communication requires timely delivery of data such as digital video and audio packets. Quality of Service (QOS) parameters specify the temporal and spatial characteristic of such continuous media data. To insure timely delivery of continuous media data, the system needs to minimize the communication delay by securing required processor and network resources. We have extended the Capacity-Based Session Reservation Protocol (CBSRP), which was proposed to realizing predictable ...

11 A beam tracing approach to acoustic modeling for interactive virtual environments Thomas Funkhouser, Ingrid Carlbom, Gary Elko, Gopal Pingali, Mohan Sondhi, Jim West July 1998 Proceedings of the 25th annual conference on Computer graphics and interactive techniques

Full text available: pdf(325.10 KB) Additional Information: full citation, references, citings, index terms

Keywords: acoustic modeling, auralization, beam tracing, spatialized sound, virtual environment systems, virtual reality

12 Analysis, modeling and generation of self-similar VBR video traffic

Mark W. Garrett, Walter Willinger

October 1994 ACM SIGCOMM Computer Communication Review, Proceedings of the conference on Communications architectures, protocols and applications,

Volume 24 Issue 4 Full text available: pdf(1.28 MB)

Additional Information: full citation, abstract, references, citings, index terms

We present a detailed statistical analysis of a 2-hour long empirical sample of VBR video. The sample was obtained by applying a simple intraframe video compression code to an action

<u>movie</u>. The main findings of our analysis are (1) the tail behavior of the marginal bandwidth <u>distribution</u> can be accurately described using "heavy-tailed" distributions (e.g., Pareto); (2) the autocorrelation of the VBR <u>video</u> sequence decays hyperbolically (equivalent to long-range dependenc ...

	and the state of t	
13	Poster session 2: GroupMedia: distributed multi-modal interfaces Anmol Madan, Ron Caneel, Alex Sandy Pentland October 2004 Proceedings of the 6th international conference on Multimodal interfaces	
	Full text available: pdf(361.16 KB) Additional Information: full citation, abstract, references, index terms	
	In this paper, we describe the GroupMedia system, which uses <u>wireless</u> wearable <u>computers</u> to measure <u>audio</u> features, head-movement, and galvanic skin response (GSR) for dyads and groups of interacting <u>people</u> . These group sensor measurements are then used to build a real-time <i>group interest index</i> . The group interest index can be used to control group displays, annotate the group discussion for later retrieval, and even to modulate and guide the group discussion itself. We explore	
	Keywords : galvanic skin response, head nodding, human behavior, influence model, interest, prosody, speech features	
14	Visual tracking for multimodal human computer interaction Jie Yang, Rainer Stiefelhagen, Uwe Meier, Alex Waibel	
	January 1998 Proceedings of the SIGCHI conference on Human factors in computing systems	
	Full text available: pdf(1.05 MB) Additional Information: full citation, references, citings, index terms	
	Keywords : face tracking, gaze tracking, lip-reading, multimodal human <u>computer</u> interaction, skin-color modeling, sound localization, visual tracking	
15	Delay jitter first-order and second-order statistical functions of general traffic on high-	
	speed multimedia networks Cathy A. Fulton, San-qi Li April 1998 IEEE/ACM Transactions on Networking (TON), Volume 6 Issue 2	
	Full text available: pdf(469.42 KB) Additional Information: full citation, references, citings, index terms	
	Keywords : QBD analysis, autocorrelation function, cell delay variation, jitter, <u>multimedia</u> , probability density function	
16	Case studies in embedded systems: The analysis and design of architecture systems for	
	speech recognition on modern handheld-computing devices Andreas Hagen, Daniel A. Connors, Bryan L. Pellom October 2003 Proceedings of the 1st IEEE/ACM/IFIP international conference on	
	Hardware/software codesign and system synthesis Full text available: pdf(280.91 KB) Additional Information: full citation, abstract, references, index terms	
	Growing demand for high performance in embedded systems is creating new opportunities to use speech recognition systems traditionally executed only on high performance systems. In several ways, the needs of embedded computing differ from those of more traditional general-purpose systems. Embedded systems have more stringent constraints on cost and power consumption that lead to <u>design</u> bottlenecks for many computationally-intensive applications.	

This paper characterizes the speech recognition pro ...

Keywords: embedded systems, performance, speech recognition 17 Simulation of nonGaussian long-range-dependent traffic using wavelets Vinay J. Ribeiro, Rudolf H. Riedi, Matthew S. Crouse, Richard G. Baraniuk May 1999 ACM SIGMETRICS Performance Evaluation Review , Proceedings of the 1999 ACM SIGMETRICS international conference on Measurement and modeling of computer systems, Volume 27 Issue 1 Additional Information: full citation, references, citings, index terms Full text available: pdf(1.54 MB) 18 Techniques for low cost spatial audio David A. Burgess December 1992 Proceedings of the 5th annual ACM symposium on User interface software and technology Full text available: pdf(727.39 KB) Additional Information: full citation, abstract, references, citings, index terms There are a variety of potential uses for interactive spatial sound in human-computer interfaces, but hardware costs have made most of these applications impractical. Recently, however, single-chip digital signal processors have made real-time spatial audio an affordable possibility for many workstations. This paper describes an efficient spatialization technique and the associated computational requirements. Issues specific to the use of spatial audio in user interfaces are addressed. The ... 19 Creating a live broadcast from a virtual environment Chris Greenhalgh, John Bowers, Graham Walker, John Wyver, Steve Benford, Ian Taylor July 1999 Proceedings of the 26th annual conference on Computer graphics and interactive techniques Full text available: pdf(1.14 MB) Additional Information: full citation, references, citings, index terms **Keywords**: multi-user, networked apps, video, viewpoint control ²⁰ Maximizing rewards for real-time applications with energy constraints Cosmin Rusu, Rami Melhem, Daniel Mossé November 2003 ACM Transactions on Embedded Computing Systems (TECS), Volume 2 Issue 4 Full text available: pdf(278.42 KB) Additional Information: full citation, abstract, references, index terms New technologies have brought about a proliferation of embedded systems, which vary from control systems to sensor networks to personal digital assistants. Many of the portable embedded devices run several applications, which typically have three constraints that need to be addressed: energy, deadline, and reward. However, many of these portable devices do not have powerful enough CPUs and batteries to run all applications within their deadlines. An

Keywords: Power management, operating systems, real-time, reward-based, scheduling

optimal scheme would allo ...

Results (page 1): (autocorrelation "auto-correlation") and microphone and coefficient and ((real t... Page 6 of 6

 $20 dsc \&short = 0 \&parser = Internet \&source_parser = Internet \&source_disp = \&source_query = \&continue to the continue to t$